

**APPARATUS FOR SYNTHESIZING SPEECH SOUNDS OF A SHORT  
MESSAGE IN A HANDS FREE KIT FOR A MOBILE PHONE**

**BACKGROUND OF THE INVENTION**

**1. Field of the Invention**

5           The present invention relates to a hands free kit for a car phone or mobile phone used in a car, and more particularly, to an apparatus for generating speech sounds of a short message received by the mobile phone by means of a digital signal processor (DSP).

**2. Description of the Related Art**

10           A hands free kit for a car phone or portable mobile phone used in a car enables the driver to safely communicate through the mobile phone without holding the phone, e.g., while both hands are on the steering wheel. (As used hereafter, the term "mobile phone" applies to both a dedicated car phone and a portable mobile phone.) The hands free kit employs a half-duplex circuit to transmit/receive audio  
15           signals to/from the handset of the mobile phone through transmit and receive lines, respectively. The half-duplex circuit functions to prevent sounds from the speaker from being inputted to the microphone, i.e., to prevent feedback.

20           A conventional hands free kit performs the communication function only when the driver has established the communication path between the handset and the hands free kit by manually operating a key on the handset. In order to eliminate such manual handset operation, the hands free kit has recently been provided with a voice control means to control the functions of the handset and to dial by voice. This feature is made possible by equipping the hands free kit with speech recognition technology embodied by a DSP therein.

Mobile phone handsets have been furnished with short message service (SMS) from communication service providers. The SMS is used to transmit a short message to the mobile phone from a caller or broadcasting service, which is displayed on the mobile phone handset display in characters. When receiving a short message, the handset outputs an audible alarm to alert the user of the reception of a message. If a short message is received while the system is operating in the hands free mode, i.e., with the handset connected to the hands free kit, the handset delivers a signal to the hands free kit indicating the same, whereupon the hands free kit sounds an alarm. The user must then manually operate the handset to read the message on the handset display. Should the user attempt to perform this task while actively driving, however, the user may create a hazardous situation.

#### **SUMMARY OF THE INVENTION**

It is an object of the present invention to provide an apparatus for synthesizing speech sounds to express a short message received in a handset coupled to a hands free kit by means of speech recognition and speech synthesis, and a method therefor.

According to one aspect of the invention, an apparatus for synthesizing speech sounds to express a short message received from a caller in a handset coupled to a hands free kit, includes handset circuitry for transferring an alarm signal to the hands free kit to generate an alarm to inform the user of the receipt of the short message. The handset transfers the short message to the hands free kit when receiving a short message calling signal from the hands free kit. The short message calling signal is generated upon the detection of a predetermined voice command by the user indicating a desire to hear the short message. The hands free kit includes circuitry for synthesizing speech sounds corresponding to the short message received from the handset.

According to another aspect of the present invention, a method for synthesizing speech sounds to express a short message in a hands free kit includes: generating an alarm upon receiving an alarm signal from the handset to inform the user of the receipt of a short message; detecting whether speech is input; detecting whether a sound data storage contains the same sound characteristics as the input speech; detecting whether the input speech is a sound synthesis command; and transmitting a short message calling signal to the handset upon detecting the sound synthesis command. The short message is then received from the handset and analyzed to synthesize the sound data corresponding to the short message by reading sound element data from a sound element code storage according to the analyzed result. The sound data is then converted into analog audio signals applied to a speaker.

The present invention will now be described more specifically with reference to the attached drawings only by way of example.

### **BRIEF DESCRIPTION OF THE DRAWINGS**

FIG. 1 is a schematic diagram of a handset connected to a hands free kit;

Fig. 2 is a block diagram of a mobile phone handset;

Fig. 3 is a block diagram of a hands free kit that synthesizes speech sounds according to an embodiment of the present invention; and

Fig. 4 is a flow chart illustrating a method of synthesizing speech sounds to express a short message in accordance with the invention.

### **DETAILED DESCRIPTION OF THE PREFERRED EMBODIMENT**

Referring to FIG. 1, a connection arrangement of a hands free kit 200, a mobile phone handset 100 and a cradle 1 is depicted. Handset 100 is mountable on

cradle 1, with the former having a hands free connector connected to hands free kit 200 through a hands free cable. The cradle 1 also typically contains electronics and a hands free kit connector. A hands free mode may be entered automatically when the handset is mounted on the cradle, or by depressing a specific key (not shown) on the handset or on the hands free kit. In the hands free mode, when a communication channel is established by receiving or transmitting a call through the handset 100, the user's speech is captured by a hands free kit microphone MIC 2 and the incoming audio wirelessly received from the other calling party is output by hands free kit speaker SP2. Hands free kit 200 transfers the audio signals received from the handset through the hands free cable to speaker SP2, or routes the audio signals generated by microphone MIC2 to the handset 100 to wirelessly transmit the same to the other calling party through an antenna ANT. The cradle 1 is also connected to the hands free kit 200 and provides signals to the hands free kit to inform the same of the state of the handset 100 mounted on the cradle 1 or the battery power level thereof. Note that antenna ANT may either be mounted on the handset itself or on the automobile; in the latter case the antenna is connected to the handset via a wire.

Referring to FIG. 2, a block diagram illustrating an exemplary configuration of handset 100 is shown. A first control unit 10 includes a read only memory (ROM) for storing an operational program, a random access memory (RAM) for temporarily storing data generated when executing the operational program, and a read/write memory such as an electronically erasable and programmable read only memory (EEPROM). An input key pad 20 includes a plurality of alphanumeric and functional keys to generate key data transferred to the first control unit 10. Control unit 10 controls a display 30 to display the key data and various information such as the operational state of the handset in icons and characters. A radio frequency (RF) part 40 modulates and demodulates the signals received and transmitted through the

antenna. A signal processor 50 processes the audio or data signals received from the RF part 40 to generate sound through a speaker SP1 or data transferred to the first control unit 10. The signal processor 50 encodes the audio signals received through the microphone MIC1, converting them into the baseband signals transferred to the RF part 40. It also converts the data received from the first control unit 10 into the baseband signals transferred to the RF part 40. The hands free connector 60 is connected through a data line to the first control unit 10, and through an audio line to the signal processor 50.

With handset 100 mounted on the cradle 1 and connected through the hands free connector 60 to the hands free kit 200, the handset 100 and hands free kit 200 are initialized according to a prescribed protocol. Then, control unit 10 controls the signal processor 50 to transfer the audio signals received from the RF part 40 to the hands free kit 200. Control unit 10 also controls the signal processor 50 to transfer the audio signals generated by the microphone MIC2 of the hands free kit 200 to the RF part 40. Generally, the hands free connector 60 of handset 100 includes a data transfer terminal and a battery recharging terminal.

With reference now to FIG. 3, a block diagram of an exemplary configuration of hands free kit 200 is shown. An interface 130, connected to handset 100 via hands free connector(s) 60 and a hands free cable (if any), separates the signals received from handset 100 into data and audio signals. A second control unit 110 includes ROM for storing an operational program, RAM for temporarily storing the data generated when executing the operational program, and an EEPROM for storing telephone numbers inputted by the user. The second control unit 110 may consist of a one-chip microprocessor including a sentence analyzer 112 to analyze the SMS short message received from the handset 100 through the interface 130.

A dictionary storage 114 stores a body of information concerning the alphabet

for the associated language including phonetic information. Storage 114 is preferably included in the EEPROM of control unit 110, but is shown separately for clarity of presentation. The sentence analyzer 112 analyzes the short message into phonetic symbols and sound elements in reference to the dictionary storage 114 so as to generate grammatical information data of the phonetic symbols and the phonetic information data of the sound elements.

A full-duplexer module 120 performs simultaneous transmission in both directions under the control of the second control unit 110 to transfer the audio signals from the handset 100 to the speaker SP2, and those from microphone MIC2 through the interface 130 to the handset 100. Module 120 includes an echo canceler (not shown) to eliminate reflective noises. A sound memory 160 includes a sound data storage 162 and a sound element code storage 164. The sound data storage 162 stores speech sound data to dial by voice in a voice dialing mode, as well as addresses in the EEPROM of stored phone numbers corresponding to the stored speech sound data. Data storage 162 also stores speech sound data corresponding to voice commands to control the functions and the addresses of the functional flags, which are stored in a specific memory of the handset 100 to set its functions. The sound code element storage 164 stores the sound element codes to represent respective alphabet letters, sub-words and words, e.g., of the Korean language.

The sound data stored within sound data storage 162 may be obtained during a set-up mode (or training mode) of the system. In particular, during the set-up mode, the user is prompted to speak one or more specific commands. One command, e.g., "read message", is an instruction for the system to proceed with a speech synthesis operation when a short message is received. The short message is then converted to synthesized or "canned" speech and output through the speaker SP2. As will be explained further below, in the hands free operating mode, when handset 100

receives a short message through the wireless communication system, an alarm is generated to inform the user of the same. Then, if the user utters the appropriate "read message" command, and the system properly recognizes the command, the system will convert the short message to audible speech. To improve the probability of successful speech recognition, during the set-up mode, the user utters the "read message" and other commands for conversion into audio feature data by a speech recognition means 140. This audio feature data is stored in sound data storage 162 as feature data to be subsequently compared against during system operation.

Additionally, the training mode is used to collect and store called party name data to be compared against in the voice dialing mode. The voice dialing mode and the short message synthesis mode are preferably concurrent modes. That is, during hands free operation, the user is able to both dial by voice and to listen to synthesized short messages.

During the hands free operating mode, speech recognition means 140 converts analog audio signals from the microphone MIC2 to sound data, which is compared with the sound data stored in sound data storage 162. When a short message is received and an alarm is generated indicating the same, the speech recognition system compares received speech with the stored command for initiating speech synthesis with respect to the message. In the voice dialing mode, the input speech is compared to the stored called party name data. In either case, when a match between input speech and stored sound data is found, speech recognition means 140 informs the second control unit 110 of the address representing the sound data of the stored speech.

The speech synthesis module 150 processes grammatical information data of phonetic symbols and phonetic information data from the sentence analyzer 112, corresponding to the SMS message, to generate sound data for conversion to analog

audio signals. This results in the received SMS message being converted to audible speech which is output by speaker SP2. Module 150 includes a control information generator 152 and a speech synthesizer 154. The control information generator 152 arranges synthetic units of the speech according to the grammatical information data of the phonetic symbols to generate the control information concerning the phonemes, pitches, strengths, lengths, tempos, rhythms, etc. of the speech sounds. The speech synthesizer 154 retrieves the sound element data from the sound element code storage 164 according to the phonetic information data to synthesize the sound data. It converts the sound data to the desired audio signals according to the control information data. The audio signals are transferred through the full-duplex module 120 to the speaker SP2. The speech recognition means 140 and sound synthesis module 150 are each preferably embodied as part of a DSP.

FIG. 4 is a flowchart illustrating a method of synthesizing speech sounds of a short message in accordance with the invention. The method steps on the left hand side, 401-405, represent steps performed in the first control unit 10 of handset 100, whereas those on the right hand side, 411 to 420, are performed in the second control unit 110 of the hands free unit. The first control unit 10 of handset 100 detects receipt of a short message in step 401. The received short message is displayed on the display 30 and stored in memory in step 402. Then, first control unit 10 informs the hands free kit 200 of the receipt of the short message by transferring an alarm signal through the hands free connector 60 in step 403.

When the alarm signal is transferred through the interface 130 of the hands free kit 200 to the full-duplex module 120 in step 403, the second control unit 110 controls the full-duplex module 120 to generate an alarm through the speaker SP2 in step 411. At this point, the system is striving to detect the previously stored, special speech command of the user, e.g., "read message". Upon proper detection of this



command, the system proceeds to convert the short message to audible speech. Simultaneously, the system is also listening for a called party name, or a previously stored telephone number, to be dialed in the voice dialing mode. If speech corresponding to a stored called party name or telephone number is recognized by means of a favorable comparison of the input speech with previously stored data in sound data storage 162, the system will first convert the corresponding stored data to an audible output to allow the user to verify the same via another voice command, prior to the system automatically dialing the telephone number.

Thus, to implement the above functions, if the second control unit 110 detects speech from microphone MIC2 in step 412, the speech recognition means 140 receives input speech from the full-duplexer module 120 to generate corresponding sound data under the control of the second control unit 110 in step 414. Then, the speech recognition means 140 compares the sound data of the input speech to ascertain whether the sound data storage 162 contains the corresponding sound data having the same (or substantially the same) sound characteristics as the speech in step 414. If so, the speech recognition means 140 retrieves the address representing the sound data to transfer it to the second control unit 110. In step 415, the second control unit 110 determines whether the data stored in the address represents a telephone number or functional command. If the data is a functional command, the second control unit 110 determines whether the functional command indicates a speech synthesis command. If it is the speech synthesis command, the second control unit 110 generates a short message calling signal transmitted through the interface 130 to the handset 100 in step 416.

Meanwhile, the first control unit 10 of the handset 100 detects receipt of the short message calling signal in step 404 after transferring the alarm signal in the previous step 403. Receiving the short message calling signal, the first control unit

10 retrieves the short message from the memory to transfer it through the hands free connector 60 to the hands free kit 200 in step 405. (It is noted here that as an alternative embodiment, the short message can be automatically transferred to the hands free kit as soon as it is received by the handset. In this case, the method steps involving generating and detecting a short message calling signal would be eliminated.)

Subsequently, the second control unit 110 of the hands free kit 200 detects receipt of the short message from the handset 100 in step 417. Receiving the short message, the second control unit 110 analyzes the sentences of the short message in reference to the dictionary of the dictionary storage 114 so as to generate the grammatical information data of the phonetic symbols and the phonetic information data of the sentences to the speech synthesis module 150. The control information generator 152 of the speech synthesis module 150 arranges synthetic units of the speech according to the grammatical information data of the phonetic symbols to generate the control information concerning the phonemes, pitches, strengths, lengths, tempos, rhythms, etc. of the speech sounds. The speech synthesizer 154 of the speech synthesis module 150 retrieves in step 419 the sound element data from the sound element code storage 164 according to the phonetic information data to synthesize the sound data. It also converts the sound data to the desired audio signals according to the control information data. In step 420, the audio signals are transferred through the full-duplexer module 120 to the speaker SP2.

While the present invention has been described with specific embodiments accompanied by the attached drawings, it will be appreciated by those skilled in the art that various changes and modifications may be made thereto without departing the gist of the present invention.